Digital Signal Processing Fundamentals (5ESC0)

## **Fourier Analysis**

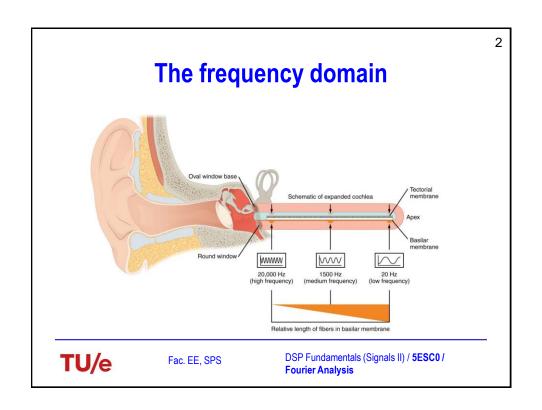
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**Notation** 

\* The book's notation:

Book :  $\omega \cdot T_s = 2\pi f \cdot \frac{1}{f_s}$ 

where  $f, \omega = \text{Absolute frequency}$ 

\* In the slides we will use:

Slides :  $\theta = 2\pi \left(\frac{f}{f_s}\right)$ ,

where  $\theta = \text{Normalized frequency}$ 

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# Fourier analysis

- \* Fourier representation of signals plays an important in both continuous-time and discrete-time signal processing
- \* It maps signals into **another "domain"** in which we can manipulate them, perform filtering
- \* Fourier representation is useful due to one of its properties: convolution operation is mapped to multiplication
- Fourier transform provides a different way to interpret signals and systems



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Continuous time Fourier transform

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- \* We will start with the continuous time Fourier transform and then continue in the digital domain
- \* Fourier Transform Continuous time signals (FTC):
- \* X(f) via  $\omega = 2\pi f$  $X(\omega) = \int_{-\infty}^{\infty} x(t)e^{-j\omega t}dt \quad \leadsto \quad x(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(\omega)e^{j\omega t}d\omega$

Example: FTC of pulse train

$$s(t) = \sum_{n = -\infty}^{\infty} \delta(t - nT) \, \leadsto \, S(\omega) = \sum_{n = -\infty}^{\infty} e^{-j\omega nT}$$



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#### **Fourier series**

- \* Only valid for periodic signals:  $x(t) = x(t + T_0)$
- If the signal is periodic, it can be described by a sum of weighted exponents

$$x(t) = \sum_{n=-\infty}^{\infty} c_n e^{j\frac{2\pi}{T_0}nt}$$

\* To find the weights  $c_n$  we look at the frequency components present in the signal and integrate them over one period and then we normalize:

$$c_n = \frac{1}{T_0} \int_{-T_0/2}^{T_0/2} x(t) \, e^{-j\frac{2\pi}{T_0}nt} dt$$



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## Fourier series example

Note: Difference FTC and FS for periodic signals

$$x(t) = \cos(2\pi F_0 t)$$
 , where  $F_o = \frac{1}{T_0}$ 

From Euler's expression we know:  $\cos(2\pi F_0 t) = \frac{1}{2}(e^{j2\pi F_0 t} + e^{-j2\pi F_0 t})$ 

$$c_{n} = \frac{1}{T_{0}} \int_{-T_{0}/2}^{T_{0}/2} \frac{1}{2} (e^{j2\pi F_{0}t} + e^{-j2\pi F_{0}t}) e^{-j2\pi F_{0}tn} dt$$

$$= \frac{1}{2T_{0}} \int_{-T_{0}/2}^{T_{0}/2} e^{j2\pi F_{0}t(1-n)} + e^{-j2\pi F_{0}t(n+1)} dt$$

$$= \frac{1}{2T_{0}} \left( \int_{-\frac{T_{0}}{2}}^{\frac{T_{0}}{2}} e^{j2\pi F_{0}t(1-n)} dt + \int_{-\frac{T_{0}}{2}}^{\frac{T_{0}}{2}} e^{-j2\pi F_{0}t(n+1)} dt \right)$$



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## Fourier series example

$$c_n = \frac{1}{2T_0} \left( \int_{-\frac{T_0}{2}}^{\frac{T_0}{2}} e^{j2\pi F_0 t(1-n)} dt + \int_{-\frac{T_0}{2}}^{\frac{T_0}{2}} e^{-j2\pi F_0 t(n+1)} dt \right)$$

We will work out one of the two integrals above, since the approach is the same.

$$\frac{1}{2T_0}(\int_{-T_0}^{T_0\over 2}e^{j2\pi F_0t(1-n)}\,dt)$$

If n=1, the exponent equals  $e^{j0}=1$  and we integrate 1 over one period, which yields  $\frac{1}{2T_0}\left(t|\frac{T_0}{2}\right)=\frac{1}{2T_0}\left(\frac{T_0}{2}-\frac{T_0}{2}\right)=\frac{1}{2}$ 



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Fourier series example

$$\frac{1}{2T_0}(\int_{-T_0}^{T_0} e^{j2\pi F_0 t(n-1)} \, dt)$$

If  $n \neq 1$ , this equals

$$\begin{split} &\frac{1}{2T_0}(\frac{1}{j2\pi F_0(n-1)}e^{j2\pi F_0t(n-1)})|\frac{T_0}{\frac{T_0}{2}}\\ &=\frac{1}{2T_0}\frac{1}{j2\pi F_0(n-1)}(e^{\frac{j2\pi\frac{1}{T_0}T_0}{2}(n-1)}-e^{-j2\pi\frac{1}{T_0}T_0})\\ &(e^{\frac{j2\pi\frac{1}{T_0}T_0}{2}(n-1)}-e^{-j2\pi\frac{1}{T_0}(n-1)})=(e^{j\pi(n-1)}-e^{-j\pi(n-1)})\\ &=\cos(\pi(n-1))+j\sin(\pi(n-1))-\cos(-\pi(n-1))-j\sin(-\pi(n-1)) \end{split}$$

Because cos(x) = cos(-x) the cosines cancel out and because n is an integer and  $sin(\pm n\pi) = 0$ , the term above is equal to 0.



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## Fourier series example

$$x(t) = \cos(2\pi F_0 t) \text{ , where } F_o = \frac{1}{T_0}$$
 
$$c_n = \frac{1}{2T_0} \left( \int_{-\frac{T_0}{2}}^{\frac{T_0}{2}} e^{j2\pi F_0 t(n-1)} dt + \int_{-\frac{T_0}{2}}^{\frac{T_0}{2}} e^{-j2\pi F_0 t(n+1)} dt \right)$$

- \* We know that the only nonzero values are at n = 1 and n = -1
- \*  $c_1 = c_{-1} = \frac{1}{2}$
- \* The frequency domain spectrum is shown in the figure below





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Special case FTC: pulse train (necessary for Ch3)

Proof of:  $S(\omega) = \sum_{n=-\infty}^{\infty} e^{-j\omega nT} \equiv P(\omega) = \frac{2\pi}{T} \sum_{k=-\infty}^{\infty} \delta(\omega - k \frac{2\pi}{T})$ 

With  $P(\omega)$  periodic with period  $\frac{2\pi}{T}$   $\Rightarrow$  we can compute the FS :

$$P(\omega) = \sum_{n=-\infty}^{\infty} p_n e^{jn\omega T}$$
, with  $p_n = \frac{T}{2\pi} \int_{-\pi/T}^{\pi/T} \frac{2\pi}{T} \delta(\omega) e^{-jTn\omega} d\omega = 1$ 

$$\Rightarrow P(\omega) = \sum_{n=-\infty}^{\infty} 1e^{jn\omega T} = \sum_{n=-\infty}^{\infty} 1e^{-jn\omega T} \equiv S(\omega)$$

⇒ FTC of pulse train

$$s(t) = \sum_{n = -\infty}^{\infty} \delta(t - nT) - S(\omega) = \frac{2\pi}{T} \sum_{n = -\infty}^{\infty} \delta(\omega - nT) - \frac{2\pi}{T}$$
 proportion

Pulse train with distance proportional to  $2 \pi /$ 

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Fourier Transform of Discretetime (FTD) signals

- In this course we work in the digital domain, therefore we will look at the Fourier Transform of discrete-time signals
- \* Because the time is discrete, the integral becomes a summation

$$X(e^{j\theta}) \stackrel{FTD}{\cong} \sum_{n=-\infty}^{\infty} x[n]e^{-jn\theta}$$

- \* Notation:
  - $\theta$  is the relative frequency and we use a capital X to denote that we are in the frequency domain
  - The only variable X depends on is  $\theta$
  - $\theta$  is continuous, so X is continuous, therefore we use round brackets
  - We could write  $X(\theta)$  because  $\theta$  is the only variable, but we write  $X(e^{j\theta})$  to stress that this is a periodic function



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# Fourier Transform of Discretetime (FTD) signals

 To go back to the discrete-time domain, we use the Inverse Fourier Transform for Discrete-time signals (IFTD)

\* Because  $X(e^{j\theta})$  is continuous and periodic, we have to integrate over one period:

$$x[n] \stackrel{\text{IFTD}}{=} \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\theta}) e^{jn\theta} d\theta$$

\* We normalize over one period with the factor  $\frac{1}{2\pi}$ 

\* Note: the Fundamental Interval (FI) is usually:  $|\theta| \le \pi$ , integrating from 0 to  $2\pi$  also works



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## Fourier Transform of Discretetime (FTD) signals

$$X(e^{j\theta}) \stackrel{FTD}{\cong} \sum_{n=-\infty}^{\infty} x[n]e^{-jn\theta} \sim x[n] \stackrel{IFTD}{\cong} \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\theta})e^{jn\theta}d\theta$$

 For common signals/ sequences we have FTD pairs that can be used without derivation

\* This saves time and possible mistakes

\* Example sequence:  $\delta[n]$ 

FTD of a delta pulse:  $\sum_{n=-\infty}^{\infty} \delta[n] e^{-jn\theta} = \delta[0] e^{-j\cdot 0\cdot \theta} = 1$ 



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**Common FTD pairs** 

 $X(e^{j\theta}) \stackrel{FTD}{\cong} \sum_{n=-\infty}^{\infty} x[n]e^{-jn\theta} \sim x[n] \stackrel{IFTD}{\cong} \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\theta})e^{jn\theta}d\theta$ 

Time Sequence	FTD
$\delta[n]$	1
$\delta[n-n_0]$	$e^{-jn_0\theta}$
1	$2\pi\delta(\theta)$
$e^{jn\theta_0}$	$2\pi\delta(\theta-\theta_0)$
$\boxed{ \qquad a^n u[n], \qquad  a  < 1}$	$\frac{1}{1 - ae^{-j\theta}}$
$-a^n u[-n-1], \qquad  a  > 1$	$\frac{1}{1 - ae^{-j\theta}}$
$\cos(n\theta_0)$	$\pi\delta(\theta+\theta_0)+\pi\delta(\theta-\theta_0)$



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#### **Example: FTD pair**

Find FTD of the sequence  $x[n] = a^n u[n]$ , |a| < 1.

The FTD of this sequence is

$$X(e^{j\theta}) = \sum_{n=0}^{\infty} a^n e^{-jn\theta} = \sum_{n=0}^{\infty} (ae^{-j\theta})^n$$

Using the geometric series, |a| < 1, this sum is

$$X(e^{j\theta}) = \frac{1}{1 - ae^{-j\theta}}.$$



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#### **FTD** properties

- \* Periodicity :  $X\left(e^{j\theta}\right) = X(e^{j\theta+l\cdot 2\pi})$   $l\in\mathbb{N}$   $X\left(e^{j\theta}\right)$  is periodic, meaning its behavior repeats every  $2\pi$
- \* Symmetry :

x[n]	$X(e^{j\theta})$
Real, even	Real, even
Real, odd	Imaginary, odd
Imaginary, even	Imaginary, even
Imaginary, odd	Real, odd

For FTD, forms of symmetry will hold as listed in the table above



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## **FTD** properties

- \* Linearity :  $ax_1[n] + bx_2[n] \leadsto aX_1(e^{j\theta}) + bX_2(e^{j\theta})$  Additive and homogeneous
- \* Shifting :  $x[n-n_0] \leadsto e^{-jn_0\theta} \cdot X(e^{j\theta})$  Shifting in time domain is a modulation operation in frequency domain
- \* Time-reversal:  $x[-n] \leadsto X(e^{-j\theta})$
- \* Modulation :  $e^{jn\theta_0} \cdot x[n] \hookrightarrow X(e^{j(\theta-\theta_0)})$ Modulation in time domain is shifting in frequency domain Example using Euler's expression:

$$\cos(n\theta_0)x[n] \sim \frac{1}{2}X(e^{j(\theta+\theta_0)}) + \frac{1}{2}X(e^{j(\theta-\theta_0)})$$



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**FTD** properties

\* Convolution : 
$$x[n] * y[n] \leadsto X(e^{j\theta}) \cdot Y(e^{j\theta})$$

\* Multiplication: 
$$x[n] \cdot y[n] \leadsto \frac{1}{2\pi} \int_{\varphi=-\pi}^{\pi} X(e^{j\varphi}) Y(e^{j(\theta-\varphi)}) d\varphi$$

Because  $X(e^{j\theta})$  and  $Y(e^{j\theta})$  are continuous, the convolution requires integration. Multiplication in one domain is convolution in the other domain.

\* Parseval : 
$$\sum_{n=-\infty}^{\infty}|x[n]|^2=\frac{1}{2\pi}\int_{\theta=-\pi}^{\pi}|X(e^{j\theta})|^2\,d\theta$$
 The energy in one domain equals the energy in the other domain; a transform does not introduce or use energy.



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#### Frequency response LTI system



- So far we had an introduction on FTD, now we will use it on an LTI system
- We know that we find the output by convolving the input with the impulse response

$$y[n] = x[n] * h[n] \stackrel{LTI}{\cong} \sum_{k=-\infty}^{\infty} x[k]h[n-k]$$

\* Now we use a complex exponent with a single frequency  $\theta$  as input:

$$x[n] = e^{jn\theta}$$



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#### Frequency response LTI system



\* Now we use a complex exponent as input:  $x[n] = e^{jn\theta}$ 

$$y[n] = e^{jn\theta} * h[n] = h[n] * e^{jn\theta} \stackrel{LTI}{=} \sum_{k=-\infty}^{\infty} h[k]e^{j(n-k)\theta}$$

\* Since the summation has index k, we can take n out of the summation

$$\sum_{k=-\infty}^{\infty} h[k] e^{j(n-k)\theta} = \left(\sum_{k=-\infty}^{\infty} h[k] e^{-jk\theta}\right) \cdot e^{jn\theta}$$

\* Now we notice a product of our input  $e^{jn\theta}$  and the part between brackets. The part between brackets does not depend on n, but only on the impulse response



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## Frequency response LTI system

$$\begin{array}{c|c}
\hline
\mathbf{z}[\mathbf{n}] \\
\hline
e^{jn\theta}
\end{array}
\begin{array}{c|c}
\hline
\mathbf{h}[\mathbf{n}] \\
\mathsf{LTI}
\end{array}
\begin{array}{c|c}
\mathbf{y}[\mathbf{n}] \\
H(e^{j\theta}) \cdot e^{jn\theta}
\end{array}$$

\* We can write the part between brackets as  $H(e^{j\theta})$ 

$$\left(\sum_{k=-\infty}^{\infty} h[k]e^{-jk\theta}\right) \cdot e^{jn\theta} = H(e^{j\theta}) \cdot e^{jn\theta}$$

- \* The relation above holds for all  $\theta$
- \* The system's response to a frequency  $\theta$  only depends on h[n]
- \* We can conclude that the impulse response h[n] and the frequency response  $H(e^{j\theta})$  are an FTD pair:

$$H(e^{j\theta}) = \sum_{k=-\infty}^{\infty} h[k]e^{-jk\theta} \quad \leadsto \quad h[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} H(e^{j\theta}) e^{jn\theta} d\theta$$



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#### **Properties frequency response**

- \* Complex:  $H(e^{j\theta}) = H_r(e^{j\theta}) + jH_I(e^{j\theta}) = |H(e^{j\theta})| \cdot e^{j\varphi(e^{j\theta})}$ We can write the frequency response as a real and an imaginary part or in polar notation (magnitude and phase)
- \* Periodicity:  $H(e^{j\theta_0}) = H(e^{j(\theta_0 + l \cdot 2\pi)})$  for  $l \in \mathbb{N}$ As denoted by  $e^{j\theta}$  in  $H(e^{j\theta})$ , the frequency response is periodic
- \* Conjugate symmetry: For real valued  $h[k] \Rightarrow H(e^{-j\theta}) = H^*(e^{j\theta})$ The magnitude is symmetric, but the phase is antisymmetric
- \* Let us consider an example:  $h[n] = \sum_{i=0}^{2} \delta[n-i]$
- \* Find  $H(e^{j\theta})$  and draw the magnitude and phase plots



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## Frequency response example

- \*  $h[n] = \delta[n] + \delta[n-1] + \delta[n-2]$
- $x^{[n]}$  T T  $y^{[n]}$
- \* We want to find the frequency response  $H(e^{j\theta})$ , which is an FTD pair with h[n]
- \* We take the FTD of h[n] and use the known FTD pair of a delayed delta pulse:  $\delta[n-n_0] \ \leadsto \ e^{-jn_0\theta}$
- \*  $h[n] \rightsquigarrow H(e^{j\theta}) = 1 + e^{-j\theta} + e^{-2j\theta}$
- \* We can take the factor  $e^{-j\theta}$  outside of brackets:  $H(e^{j\theta}) = 1 + e^{-j\theta} + e^{-2j\theta} = e^{-j\theta}(e^{j\theta} + 1 + e^{-j\theta})$
- \* We can rewrite this to use an Euler expression:

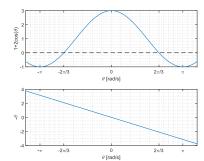
$$e^{-j\theta}\left(e^{j\theta}+1+e^{-j\theta}\right)=e^{-j\theta}\left(1+2\left(\frac{e^{j\theta}+e^{-j\theta}}{2}\right)\right)$$



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#### Frequency response example

- \*  $h[n] = \sum_{i=0}^{2} \delta[n-i]$
- \*  $H(e^{j\theta}) = e^{-j\theta} \left( 1 + 2\left(\frac{e^{j\theta} + e^{-j\theta}}{2}\right) \right) = (1 + 2\cos\theta)e^{-j\theta}$
- \* From the complex property we know that  $(1+2\cos\theta)$  describes the magnitude and  $-\theta$  describes the phase
- \* Now we plot the magnitude and phase as a function of  $\theta$



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## Frequency response example

- \*  $(1 + 2\cos\theta)$  describes the magnitude and  $-\theta$  describes the phase
- \* In practice, both the magnitude and phase are plotted within the fundamental interval:  $|\theta| \leq \pi$
- \* The magnitude  $|H(e^{j\theta})|$  is often taken absolute and the phase  $\angle\{H(e^{j\theta})\}$  is limited from  $-\pi$  to  $\pi$
- \* When taken absolute, where the magnitude would cross zero the value is made positive. In other words: a phase shift of  $\pi$ . The phase plot follows this behavior by a  $\pi$  phase shift
- \* In the phase plot,  $\pi$  is added or subtracted so that the phase stays within the limits of  $-\pi$  to  $\pi$  when the magnitude plot crosses zero



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#### **Properties frequency response**

\* Solving DE:

$$y[n] = -\sum_{k=1}^{p} a_k y[n-k] + \sum_{k=0}^{q} b_k x[n-k]$$

$$\sim Y(e^{j\theta}) = -\sum_{k=1}^{p} a_k e^{-jk\theta} Y(e^{j\theta}) + \sum_{k=0}^{q} b_k e^{-jk\theta} X(e^{j\theta})$$

$$H(e^{j\theta}) = \frac{Y(e^{j\theta})}{X(e^{j\theta})} = \frac{\sum_{k=0}^{q} b_k e^{-jk\theta}}{1 + \sum_{k=1}^{p} a_k e^{-jk\theta}}$$

- A difference equation in frequency domain can be solved with linear algebra
- \* Convolution:  $x[n] * h[n] \rightsquigarrow X(e^{j\theta}) \cdot H(e^{j\theta})$



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#### **Example FTD**

Find the FTD of  $x[n] = a^n \sin(n\theta_0) u[n]$ 

#### Solution

Express the sinusoid as a sum of two complex numbers using *Euler's* formula:

$$\sin(n\theta_0) = \frac{1}{2i} \left( e^{jn\theta_0} - e^{-jn\theta_0} \right)$$

Therefore, the expression becomes:

$$x[n] = a^{n} \frac{1}{2j} (e^{jn\theta_{0}} - e^{-jn\theta_{0}}) u[n]$$
  
=  $\frac{1}{2j} a^{n} e^{jn\theta_{0}} u[n] - \frac{1}{2j} a^{n} e^{-jn\theta_{0}} u[n]$ 



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#### **Example FTD**

Let's take the FTD of the first term which is equal to  $\frac{1}{2i}a^ne^{jn\theta_0}u[n]$ :

$$X_1(e^{j\theta}) = \frac{1}{2i} \sum_{n=0}^{\infty} a^n e^{jn\theta_0} e^{-jn\theta}$$

Take the common power n outside the bracket and join the powers of eover a single exponential

$$X_1(e^{j\theta}) = \frac{1}{2j} \sum_{n=0}^{\infty} (ae^{-j(\theta-\theta_0)})^n$$

Use the geometric series definition for  $\alpha^n u[n]$  where  $\alpha=ae^{-j(\theta-\theta_0)}$   $X_1\Big(e^{j\theta}\Big)=\frac{1}{2j}\frac{1}{1-ae^{-j(\theta-\theta_0)}}$ 

$$X_1(e^{j\theta}) = \frac{1}{2j} \frac{1}{1 - ae^{-j(\theta - \theta_0)}}$$



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## **Example FTD**

Take the FTD of the second term using the same steps:

$$X_{2}(e^{j\theta}) = -\frac{1}{2j} \sum_{n=0}^{\infty} a^{n} e^{-jn\theta_{0}} e^{-jn\theta}$$
$$= -\frac{1}{2j} \sum_{n=0}^{\infty} (ae^{-j(\theta+\theta_{0})})^{n}$$
$$= -\frac{1}{2j} \frac{1}{1 - ae^{-j(\theta+\theta_{0})}}$$

The final expression for  $X(e^{j\theta}) = X_1(e^{j\theta}) + X_2(e^{j\theta})$  is:

$$X(e^{j\theta}) = \frac{1}{2j} \left( \frac{1}{1 - ae^{-j(\theta - \theta_0)}} - \frac{1}{1 - ae^{-j(\theta + \theta_0)}} \right)$$



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#### **Filters**

- Digital filter is often used to refer to a discrete-time system
- \* A definition¹ of digital filter "...computational process or algorithm by which a sampled signal or sequence of numbers (acting as the input) is transformed into a second sequence of numbers termed the output signal. The computational process may be that of lowpass filtering (smoothing), bandpass filtering, interpolation, the generation of derivatives, etc."
- \* Filters may be characterized in terms of their system properties, such as linearity, shift-invariance, causality, stability, etc.

 $<sup>^{\</sup>rm 1}$  System Analysis by Digital Computer. F. F. Kuo and J. F. Kaiser, Eds.. John Wiley and Sons, New York. 1966



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## **Examples: Averaging filter**

Time domain	00	Frequency domain
Averaging filter		Dirichlet function
h[n] = u[n] - u[n - N]	0-0	$H(e^{j\theta}) = e^{-j\frac{N-1}{2}\theta} \cdot \frac{\sin(\frac{N}{2}\theta)}{\sin(\frac{1}{2}\theta)}$

- \* The values of the impulse response h[n] should be multiplied by 1/N to obtain the averaging effect
- \* An array of *N* delta pulses can be described in the frequency domain by the Dirichlet function
- We will look at the derivation and the magnitude plot



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#### **Examples: Averaging filter**

We take a factor  $\frac{2je^{-j\theta N/2}}{2je^{-j\theta/2}}$  out of brackets to use an Euler equation:

$$\frac{1 - e^{-j\theta N}}{1 - e^{-j\theta}} = \frac{\left(\frac{e^{-\frac{j\theta N}{2}} - e^{\frac{j\theta N}{2}}}{2j}\right)}{\left(\frac{e^{j\theta/2} - e^{-j\theta/2}}{2j}\right)} \cdot \frac{2je^{-\frac{j\theta N}{2}}}{2je^{-\frac{j\theta}{2}}} = \frac{\sin(\frac{\theta N}{2})}{\sin(\frac{\theta}{2})} \cdot e^{-\frac{j\theta N}{2} + \frac{j\theta}{2}}$$
$$= \frac{\sin(\frac{\theta N}{2})}{\sin(\frac{\theta}{2})} \cdot e^{-j\theta(\frac{N-1}{2})}$$

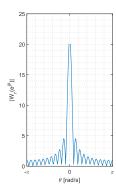


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## **Examples: Averaging filter**



- There are many zero crossings in the magnitude (not in the figure because it is taken absolute)
- The main lobe of the magnitude is quite narrow
- Our time domain block (array of delta pulses) has length 21
- Stretching in one domain is shrinking in the other domain

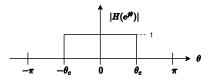
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## **Examples: Low Pass Filter**

Time domain	00	Frequency domain
Sinc function $\frac{\theta_c}{\pi} \left( \frac{\sin(\theta_c n)}{\theta_c n} \right)$	<b>⊶</b>	Ideal Low Pass Filter (LPF) $H(e^{j\theta}) = \begin{cases} 1, &  \theta  \le \theta_c \\ 0, & \text{elsewhere} \end{cases}$

- \* If we want to filter out higher frequencies, we use a LPF
- \* An ideal Low Pass Filter is shown in the figure below



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#### **Examples: Low Pass Filter**

 We can obtain the time-domain function that will give us a Low Pass Filter through the IFTD

$$\begin{split} H\!\left(e^{j\theta}\right) &= \begin{cases} 1, & |\theta| \leq \theta_c \\ 0, & \text{elsewhere} \end{cases} \\ h[n] &= \frac{1}{2\pi} \int_{-\pi}^{\pi} \! H\!\left(e^{j\theta}\right) e^{jn\theta} d\theta = \frac{1}{2\pi} \int_{-\theta_c}^{\theta_c} \! 1 \, e^{jn\theta} d\theta \\ &= \frac{1}{2\pi} \! \left(\frac{1}{jn} e^{jn\theta}\right)_{-\theta_c}^{\theta_c} = \frac{1}{\pi n} \cdot \frac{1}{2j} \! \left(e^{jn\theta_c} - e^{-jn\theta_c}\right) = \frac{\sin(n\theta_c)}{\pi n} \end{split}$$

This can be written as:  $h[n] = \frac{\theta_c}{\pi} \cdot \frac{\sin(n\theta_c)}{n\theta_c}$ , because  $\operatorname{sinc}(x) = \frac{\sin(x)}{x}$ 



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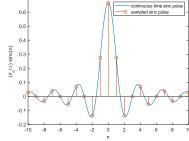
**Examples: Low Pass Filter** 

\* So the time-domain function that will give us a Low Pass Filter in frequency domain is a sinc function:  $h[n] = \frac{\theta_c}{\pi} \cdot \frac{\sin(n\theta_c)}{n\theta_c}$ 

\* The sinc function spans from  $-\infty$  to  $\infty$ , so it is infinite and we cannot make it in practice sinc function for -10  $\le$  n  $\le$  10, with  $\theta_c$  =2 $\pi$ /3

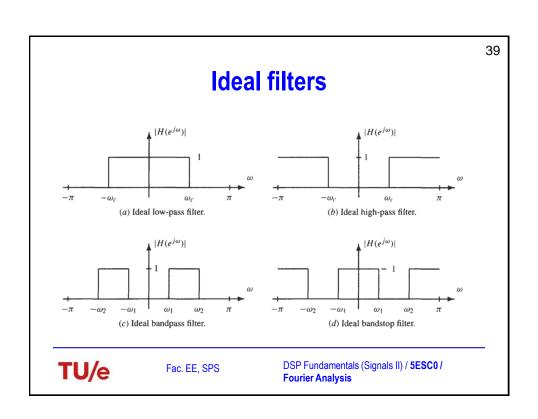
\* This means the LPF will not be ideal: there will be a ripple and the transition will not be a straight line but a slope.

 On the right, a sinc pulse and the sampled version are shown



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## **Example: windowing**

- \* We cannot make infinitely long sequences/ signals. What will the spectrum of these signals look like in practice?
- \* We will consider an example:  $x[n] = \cos(0.28\pi n)$
- We know from the Fourier Series that this will result in a spectrum of two weighted delta pulses
- \* The cosine function spans infinitely long, so in practice we cannot generate it. We can only generate a part of it



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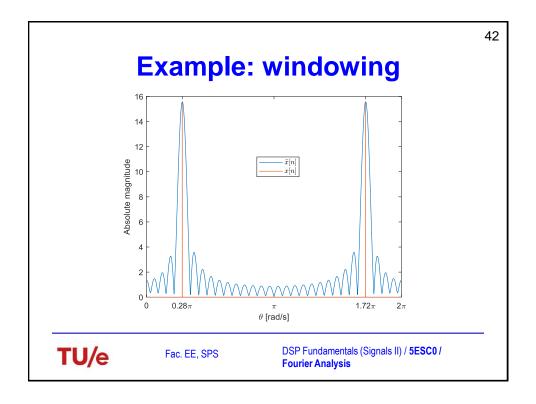
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## **Example: windowing**

- $* x[n] = \cos(0.28\pi n)$
- \* To make it finite, we multiply x[n] with a weight function (or a rectangular window)  $w_R[n] = \begin{cases} 1, & n = 0, ..., N-1 \\ 0, & \text{elsewhere} \end{cases}$
- \*  $w_R[n]$  is an averaging filter, as we have seen before
- \* Say we name the signal  $\tilde{x}[n] = x[n] \cdot w_R[n]$
- Now we observe what happens when we use the FTD on both signals

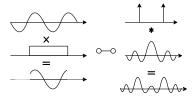


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# **Example: windowing**

- \* Why do we see this behavior?
- \* We multiply the time domain cosine with an averaging filter that gives us only the windowed part of the cosine
- \* In frequency domain this means that we have two delta pulses that are convolved with the Dirichlet function
- Convolving with a delta pulse shifts (modulates) the function you are convolving with to the position of the delta pulse



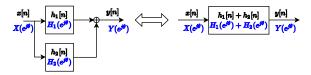
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#### Interconnection of systems

- Frequency response properties of interconnecting systems
- We can describe interconnecting systems as one system
- Cascaded: convolution in time domain is multiplication in frequency domain
- Parallel: addition is the same in both domains







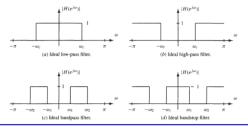
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## **Example: system connections**

- \* The cascade of a low-pass filter with a high-pass filter may be used to implement a bandpass filter.
- For example, the ideal bandpass filter shown in (c) may be realized by cascading a low-pass filter with a high-pass filter that has lower cutoff frequency
- \* Similarly, the bandstop filter shown in (d) may be realized with a parallel connection of a low-pass filter and a high-pass filter





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#### **Summary**

- We had an introduction on the Fourier Transform of Continuous time signals (FTC).
- \* We considered the Fourier Transform of Discrete-time signals (FTD)
  - FTD pairs
  - FTD properties
- \* We introduced the frequency response
  - The system's response to an input signal  $e^{jn\theta}$
  - It forms an FTD pair with the impulse response
  - Frequency response properties
- We looked at some filter examples in time domain and frequency domain: the averaging filter and the low pass filter
- \* We saw an example of what happens when we window a signal
- We examined interconnection of systems in time domain and frequency domain



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#### Reference

M. H. Hayes, "Schaum's Outline of Theory and Problems of Digital Signal Processing", McGraw-Hill, 1999; Chapter 2.



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